

APPLICATION  
FOR  
UNITED STATES LETTERS PATENT

TITLE: LIGHTWEIGHT INTERNET PROTOCOL TELEPHONY  
CLIENT

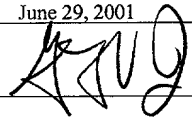
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**LIGHTWEIGHT INTERNET PROTOCOL TELEPHONY CLIENT**Background

[0001] The present application describes systems and techniques relating to Internet Protocol (IP) telephony and more particularly to a lightweight client application for IP telephony.

[0002] IP telephony enables allows callers to use a packet-switched network, such as the Internet, as the transmission medium for making telephone calls. As shown in Fig. 1, an IP telephony client application 101 (a software application program executing on a computer platform such as a PC) converts a user's voice signals into digital format, encapsulates the digitized voice data into IP packets, which are then transmitted via a communication link 100 to a network 103, such as a LAN (local area network) or WAN (wide area network). Depending on the identity of the called party, the IP packets containing the digitized voice data can be delivered either to another client application 102 connected to the network 103, or to a conventional telephone handset 106 connected to the Public Switched Telephone Network (PSTN) 105 via a gateway 104.

Drawing Descriptions

[0003] Fig. 1 is a block diagram of a typical network configuration that supports IP telephony.



feedback to the user regarding information such as the number being dialed, call status, and the like.

[0010] Prior to using an IP telephony client, a user typically first must visit a website associated with a client vendor or other software distributor, download the client application (usually about a megabyte or larger in size) to the user's computer, and then install the client application. This procedure typically results in a static copy of the client application residing on the user's computer system. Once installed, the user can make calls by executing and interacting with the IP telephony client, which in turn typically communicates with a IP telephony server residing elsewhere in the network using a proprietary (e.g., non-standard) protocol to provide IP telephony service.

[0011] Fig. 3 is a block diagram of an IP telephony network configuration 300 that implements a "stimulus"-based IP telephony client 310, residing on a user's computer platform, in communication with a feature server 304 to provide enhanced IP telephony functionality. As used herein, a stimulus client may be one that provides very little, if any, functionality locally but rather passes through input from a user to a remote server, which in turn provides the requested functionality. The stimulus client may communicate with the server by means of "stimulus signaling." Put another way, a stimulus client may be one that acts as a stimulus that causes the server

to provide functionality requested by a user at the client.

By moving the software infrastructure that provides the bulk of the functionality to the server side, a stimulus client can be made "lightweight" - that is, very small in size - while still providing the user with a rich and robust feature set.

[0012] As shown in Fig. 3, the stimulus client 310 and the feature server 304 reside within a packet-based network 302. The feature server 304 is a functional entity that uses stimulus signaling to provide an endpoint with additional features not available locally at the endpoint.

An example of a feature server implementation is described in International Telecommunication Union (ITU-T) H.323 Annex L: Packet-Based Multimedia Communications Systems (March 2001).

[0013] The feature server 304 can communicate not only with the stimulus client 310 but also with other IP telephony endpoints within the packet-based network (e.g., terminal 308 or terminal 312) and/or via gateway 306 with endpoints such as telephone 316 in the PSTN 314. The feature server 304 is able to communicate with endpoints using any of various standard call control protocols including the ITU-T H.323 protocol, the Session Initiation Protocol (SIP) as defined in Request For Comment (RFC) number 2543 (Mar. 1999) of the Internet Engineering Task Force (IETF), the Media Gateway Control Protocol (MGCP) as defined in IETF RFC number 2705 (Oct. 1999), and the ITU-T





the stimulus client's stack (the software code, typically implemented as dynamically linked libraries (DLLs), that are invoked to provide telephony services) can be kept to a bare minimum. A small client stack size may be desirable both to client vendors and to end-users - client vendors tend to like it because a small stack generally simplifies development, maintenance and distribution of client applications and end-users tend to like it because of dramatically reduced client download times. Indeed, as a result of its small size, downloading of the stimulus client can appear to end-users as being virtually instantaneous. In contrast, a download of a large-stack client - typically about 1 megabytes or larger - can take several minutes, especially when the user has a standard telephone-line connection that is limited to 56 Kbps or slower. Consequently, users are likely to be much less resistant to downloading the stimulus client because doing so entails little or no waiting time.

[0017] Moreover, the lightweight client may be advantageous from the perspective of client vendors because, in view of the minimal download time, users are much more likely to download and use the most current version of the client. In fact, use of a stimulus client coupled to a feature server can reduce the size of the client to an extent that it becomes practical for the client to be downloaded dynamically, and transparently from the user's perspective, with each new instantiation. In



that case, each user is guaranteed to use the most recent version of the client. Consequently, software development and maintenance costs can be reduced dramatically because a lightweight client vendor would no longer have to worry about version control or compatibility with earlier versions. At the same time, because each use of the client may involve a dynamic download and instantiation of the client, client vendors can more easily and continuously (or frequently) improve or upgrade the client to provide users with cutting-edge functionality. Further, dynamically downloading the client with each use allows client vendors to more easily keep track of user and usage statistics.

[0018] Other advantages may arise as a result of the stimulus client's support and use of standard IP telephony protocols such as MGCP, H.248, SIP and H.323. For example, in contrast to a conventional IP telephony client, which typically communicates with the server using a proprietary protocol, the stimulus client can communicate with the feature server using MGCP or H.248. The feature server in turn can communicate with other endpoints and network entities using SIP and/or H.323. Consequently, the stimulus client and the feature server are able to interoperate with any IP telephony endpoint or service provider that uses standard protocols. For example, by collecting DTMF data from an end-user and passing it on to the feature server, the stimulus client can provide the

end-user with a full-range of supplementary services such as defined in H.450.

[0019] Fig. 4 is a block diagram showing details of a stimulus client configuration 400 in which the stimulus client 402 communicates with a call agent 404 using a standard IP telephony protocol (e.g., MGCP or H.248) over communication link 411. The stimulus client 402 has two basic components: an application layer 406, which represents the software layer that interacts with the end-user to present a user interface, collect DTMF information and the like, and a MGCP stack 410, which is the set of software routines that facilitates MGCP-based communications with the call agent 404. The application layer 406 communicates with the MGCP stack 410 through a plugable call control (PCC) application program interface (API) 408. The PCC API 408 allows a single client application to place calls through multiple protocols utilizing a single, common API.

[0020] More particularly, the PCC API 408 exposes a common set of function calls, properties, and callbacks that can be used with any of several different call control protocols such as MGCP, H.248, SIP and H.323. It is preferable that the PCC API 408 exposes the fewest possible number of parameters for each function call so as to provide the simplest usage model for the default call model. Examples of common function calls, properties, and callbacks that may be exposed by PCC API 408 include (1)

listening for incoming calls, (2) placing calls, (3) answering calls, (4) hanging up calls, (5) capability selection, negotiation and renegotiation, (6) security (authentication, integrity), (7) call hold, (8) mute, (9) establishing multi-point conferences, merging multiple conferences into one, (10) splitting a conference into two, (11) transferring with and without consultation, (12) overlapped sending and dialing, (13) sending DTMF signals, (14) multi-line/multi-call/multi-station appearance, (15) park/pickup calls, (16) redirect/forward calls, and (17) out-of-band service commands.

[0021] The call agent 404 is a server-side application that includes a feature server 403, a signaling gateway 405, one or more stacks 412-414 for communicating with endpoints such as stimulus client 402 and/or called endpoint 416, and a PCC API 408 for each different stack instantiation 412-414. The signaling gateway 405 serves as a signaling front-end to the feature server 403 to enable the feature server 403 to communicate with endpoints using different signaling protocols such as SIP, MGCP, H.248 and H.323. As shown in Fig. 4, the call agent 404 includes a separate MGCP stack 412 for communicating with MGCP-based endpoints such as the stimulus client 402. Similarly, the call agent 404 would include a separate stack 414 (e.g., SIP, H.323, H.248 or MGCP) for each different protocol type that is to be used.



First, a user accesses a web browser application and visits a URL (uniform resource locator) address associated with an IP telephony website (e.g., an IP telephony client vendor or service provider website) (502). Next, input received from a user, for example, by clicking a button or a link, initiates an instance of an IP telephony client (504). In other words, the user gives an indication by mouse click or otherwise that usage of the IP telephony client is desired.

[0026] Alternatively, the user could have previously downloaded and stored on the computer platform locally a small, dedicated application whose primary, or sole, purpose is to download the most recent version of the IP telephony client stored at a specified URL. In that case, the user would not need to access a web browser to use the client, but rather could, for example, simply double-click on a desktop icon corresponding to the dedicated download application.

[0027] In response to the user's indication that usage of the IP telephony client is desired, the current version of the IP telephony client downloads from the website and launches (506), for example, presenting the user with a user interface similar to that shown in Fig. 2. The client can be implemented as a Java applet that downloads and is executed locally by a virtual machine (VM) on the user's computer platform. Other implementations of the client can use interpreted or scripting languages such as PERL, TCL or



[0030] Other embodiments may be within the scope of the following claims.

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